

FREE 96 PAGE AMIGA MUSIC GUIDE

Doctor Horgan's Complete Guide to

Amiga Music and FX



All you need to know about:

- Musical uses for your Amiga • The Amiga's internal music
- Perfect sampling • Sequencer techniques • MIDI & recording extras
- Amiga samplers • Creating sound effects & much, much more ...

AMIGA
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From the *Amiga Doctor Show*

Amiga Music and FX



By Tony Hargan
(Dedicated to Jo, Patroness of Patience)

Designed by Anthony Collins

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The Amiga's audio hardware and MIDI facilities



Meet Paula

Let me introduce you to Paula. Paula is the chip inside your Amiga that is responsible for every sound your Amiga makes. Paula is not an acronym, all of the Amiga's custom chips were given names when they were originally developed so if you want to look inside your Amiga you'd also find chips called Gary, Dennis, Agnes and even the rather unflattering, Fat Agnes. Paula deals with a few tasks, including the control of the disk drives but its main job as far as we are concerned is to play sound and music.

8-bit architecture

Unlike the rest of the Amiga's architecture, Paula is an 8-bit chip. This means that the highest number it can receive around is a single factor of 256 (much of

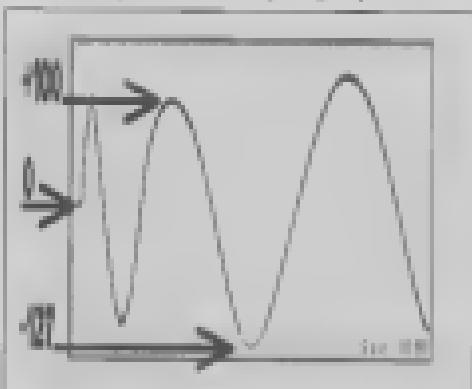


Working with these samples is quick and easy. Data is reported for the assay using cutting edge technology to provide results in a timely fashion.

the components of the current Atmos range has 24-bit architecture, which means that the highest number these parts can deal with is 4.295 million. Why should we care about that? Because the 8-bit architecture directly relates to the maximum bandwidth of the sounds that Peavey can output. Most modern digital musical instruments have 16-bit architecture or better. You would have to work quite hard to get bad sound quality with a 16-bit system, but with an 8-bit system the reverse is true. Higher bandwidth leads to higher fidelity of sound. Don't worry though, this doesn't mean you're stuck with useless technology! Far from it in fact. Many people make a living from producing music within the confines of 8 bits, but you do need to know what you're doing, in order to get those professional results and that is what this book is for.

Sampled sources

The Amiga generally deals with sampled sounds. Although Proela can play sounds that have been synthesised (sound waves created by mathematics), it formulates these tend to be rather uninteresting and cannot compete with even the most basic synthesiser or keyboard. However, using sampled sounds you can create absolutely any sound you like. Sampling is a term used to describe the process of making a digital recording of a sound. Using a sampler cartridge you can sample any sound and then replay it from your Amiga. During the sampling process, the sampler cartridge reads the incoming sound wave thousands of times every second. Each time it reads the wave it records a value between 0 and 255 (or 128 to +128) depending on the position of the wave at that precise moment. By recording a series of these snapshots, it builds up a digital picture of the



音節を複数の音節に分けて、各音節を発音する。この音節を複数の音節に分けて、各音節を発音する。

signed with. The Amiga can then replay the sound by moving through each of these snapshots very quickly. This process gives the impression of a moving sound wave in the same way that a series of still frames of a motion film give the illusion of moving pictures. The ear is fooled into thinking that it can hear the original sound.

Sample rates

The rate at which these snapshots are taken has a direct effect on the quality of the sound recording. (The sampled sound). If the rate at which the snapshots are taken is too low the sample will not sound exactly the same as the original sound. To use the camera/lens analogy once more, imagine if you were watching a film and the picture only changed once every second instead of 30 times every second.



Showing the eight samples ready when recording a sound or musical file. You can also make a new representation of the original sound.

All the movement would look very jerky and the illusion would be ruined. The same is true of a sampled sound - in this case the result would be audible footsteps crossed by the steps of the sound wave. Each step would make a slight clicking sound and together the effect of these clicks would be an unpleasant ringing tone while certain parts of the original sound would simply not have been recorded - particularly the highest frequencies.

For this reason sample rates are very important when it comes to sound quality. Paul can replay sampled sounds at a maximum rate of 22kHz, which means 22,000 snapshots every second. These incredibly high rates are required to replay the high frequencies that the human ear can detect. This also allows you to replay sounds at high pitches that were originally recorded at lower rates which is essential for musical applications.

Four Channels

Four mono channels of sound can be played simultaneously. Alternatively you can play two stereo channels. In effect it's the same thing. All of the sound from Paul is passed through the two RCA phono sockets at the back of your Amiga. If you are using four mono tracks (as is the case with most music packages) then two of the tracks will be output to one of the RCA phono sockets and the other two will be passed to the remaining phono socket. In this case if you were to connect these audio outputs to a hi-fi (using an A/B, or L/R input on the rear of the fi-fi for instance) you would then have two of the channels through the left speaker and the other two from the right speaker.

Trackers work with four channels of mono samples, while MIDI-generated sequences that have Amiga sample libraries often work with mono sam-

play but choose the output channel at random. The more refined sequencers allow you to specify the output channel for a sample. For mixed purposes it is naturally best to work with stereo samples as this allows you to play four sounds at once (eg bass, drums, melody and vox). If you are using stereo samples you are limited to playing just two at once. This is because a stereo sample is actually two mono samples (one for the left and one for the right). So by playing two stereo samples you are using up all four output channels. Stereo samples are most often used in multimedia type applications where samples may be replayed one at a time. For example, *VocalStage Pro* (which came free with the March 1990 issue of *CU Amiga Magazine*) can load and play stereo samples within presentations. Most sampler cartridges offer stereo sampling, and their accompanying software will often have facilities to replay a series of stereo samples in the form of a playlist. *AudioMaster IV* (free with the November 1990 issue of *CU Amiga Magazine*) has advanced stereo replay facilities which even allow you to create sequences from a single sample by setting up various loop points. Using this, with enough memory you could sample an entire song and completely reconstruct it by remapping sections and loops.

Chip RAM

Your Amiga has two types of memory: Chip RAM and Fast RAM. Chip RAM has a very special relationship with Paula as it does with all the Amiga's custom chips. As standard, Paula can only play samples that are stored in Chip RAM, although now many samplers and trackers can play samples from Fast RAM. Chip RAM is sometimes called Graphics Memory because it is also the memory that is used to display graphics. Old 1.3 Amiga 500s have 8.5Mb

of Chip RAM, while the A500 Plus and A600 both have 16Mb of Chip RAM. The A1200 and A4000 both come with 20Mb of Chip RAM. Most RAM expanders will add to the total Fast RAM of the Amiga but not affect the Chip RAM capacity.

However, there are Chip RAM upgrades available such as the MegaChip RAM £1599 from Power Computing which upgrades an A300 or A2000 to 2Mb of Chip RAM. Other Amigas can also be upgraded at a similar cost with a replacement Agnus chip £299 also available from Power Computing. Chip RAM limitations are becoming less of a problem as programmers develop ways to play samples from Fast RAM.

Amiga MIDI

The other side of Amiga music is MIDI sequencing.



One of the many popular MIDI sequencers on the Amiga is shown at the editor stage. It has also had a powerful sequencer which makes it very versatile.

MIDI sequencing is the term used to describe a computer (or other device) mainly controlling any number of electronic musical instruments so that they produce music. This is a totally separate area from Amiga sampling, but the two can work together very neatly. MIDI stands for Musical Instrument Digital Interface and is a standard form of communication used by all modern electronic musical instruments.

MIDI instruments talk to each other via standard MIDI cables. These do not carry audio signals but are data lines. MIDI cables have a five pin DIN plug at each end, but there are no such connectors on the Amiga. However, all you need is a MIDI interface, plugged into the Amiga's serial port to be able to connect directly to any MIDI instrument. MIDI instruments include drum machines, keyboards, synthesizers, synthesizer modules, effects

processors, outboard samplers and even mixers which can be programmed to have their faders move up and down at the appropriate point in a track.

A typical basic MIDI set-up might consist of an Amiga connected to a synthesised keyboard via a MIDI interface and a couple of MIDI leads. MIDI leads have a one-way flow of information, so one is needed to carry information to the keyboard from the computer while another is needed to take information back from the keyboard to the computer. The Amiga would be running some MIDI sequencing software. This software would be able to record any music that was played on the keyboard. However, as the lead is a data link, the computer would only record whatever notes were played, for how long, how hard they were hit and so on. No audio recording is made. For example the message travelling down the MIDI cable from the keyboard to the computer (if translated into English) could be something like "C 2 played softly for half a second, notes C 3 F 3 and A 4 all played at once for two seconds..." When the computer had finished recording the music could be played back by sending that same information back along the MIDI cable to the keyboard, which would then play the notes as it was told to do so by the computer.

From within the sequencer software running on the Amiga you could examine and rearrange the notes, correcting any mistakes you might have made in the timing (for example, frequencies all come with an optional 'quantise' function which will automatically correct the timing of all the notes according to its own internal mathematics). Recording the order of the notes and so on rather than the actual sound has many advantages. For example you can record each track separately (drums first,



Here we have a MIDI sequencer which lets you play out the various instruments in sequence. All in Amiga software and you can do it too!

baseline next, then drums and melody etc) and if you're not happy about one particular part you can always change or delete it without affecting the rest of the arrangement. It also gives you the benefits of a multi-track tape recorder without the expense.

You can do all the mixing from within the mixer, setting the relative volumes of each track and so on, recording the whole thing to a single stereo tape or DAT recorder.

As you'll see in the Sequencers and Trackers section of this book, it's quite possible to combine Amiga sampled and MIDI sequencing. In fact this is the best option if you really want to use your Amiga's full musical potential.

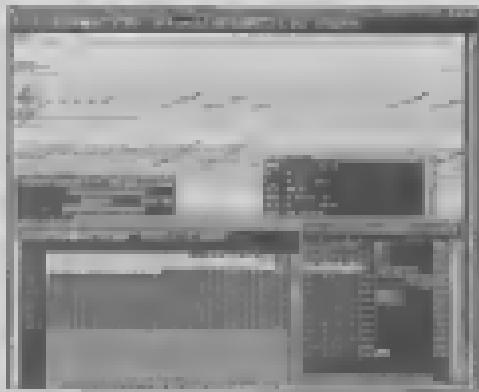
Musical Applications For Your Amiga



The obvious role for your Amiga is to fill in a musical situation as that of a sequencer: playing music using either internal samples, MIDI instruments or a combination of both. However, there are plenty of other uses to which you can put your Amiga in the studio. This chapter will highlight some of the uses and reveal where that your Amiga can lend a hand in the music production process.

Sequencer

As the subject of sequencers and trackers is explored quite thoroughly in Chapter 8 we won't go into too much detail here. Running a tracker or sequencer program on your Amiga gives it the power to control almost the entire music production process, from sampling sounds on the fly, to composing and even performing an automated musical



With a digital audio workstation you can record in stereo and produce the highest quality music from your home studio.

driven to your recorder. In this case, the Amiga is the master of the studio, while the rest of the equipment are slaves.

Digital mastering

Digital audio recording has revolutionised music production over the last few years. Until recently if you wanted to make a high fidelity master recording of a piece of music, the chances are you would have to buy an expensive studio with a professional reel to reel tape recorder with the additional cost of having an engineer to path all the right buttons. These days almost everything is recorded digitally. Now that CD is firmly established as the international standard for commercial music distribution, there's even that reason to go straight to digital at the mastering stage.

Your Amiga has the potential to act as a professional quality digital recorder, making recordings

that match the standard of DAT and CD. CDs are recorded digitally at 44.1KHz with Delta modulation. By adding a Tascam or Samson AD816 card to your Amiga it can record audio straight to a hard drive at exactly the same rate. This is a major benefit as it means that in a perfect scenario you can make one master recording from your own home studio and transfer that in a single digital to digital copy onto a CD. This means you get the cleanest possible recording, as there is only one point at which unwanted noise and distortion can degrade the recording, the point at which the music is mixed and fed into the recorder.

The elimination of noise and distortion is not the only advantage of using the Amiga as a digital recorder. Once you have the music on your hard drive you are then free to make any edits you like. The music will have been saved at an extremely large



With a digital audio workstation you can make of quality recordings in your home studio that are not bad ones.

sample, and this can be loaded into memory where it can be cut, pasted, cleaned up and passed through a number of effects using specialised software. For example, if you had originally recorded a ten minute tape but radio stations wanted a shorter version for play, copying the recording and snipping out bits here and there would be a very simple task. For the second copy you were making, was made in a straight digital to digital process there would be no loss of sound quality as there would be if you had to copy an original tape recording to another tape for musical spacing. In fact you could make as many copies of the original as you needed, editing such as a specific audience (album version, radio friendly edit, dance mix etc.)

Acoustic recording

Even an extremely simple (cramped) set up would benefit greatly from a digital recorder. Cards such as Tascam and Samson AD916 will accept any audio signal input as a digital file. If you're recording electric or bass guitar in many ways the advantages of digital recording and editing are more numerous for non-MIDI set-ups, where instruments are recorded live.

Let's say you were recording a bassline onto a track using an electric bass guitar. Halfway through the track your bass goes out of the window and you lose it completely, although you pulled off an extremely disastrous stop fill during the intro passage. If you were recording to analog tape, you would have two options: either start the whole bassline recording again and lose that initial pick, of guitar or attempt to drag on half way through the track and resume recording from where you started up, which can be tricky. Neither is a particularly attractive proposition, but if you were recording to a hard drive you could carry on.



Load up your 2000 samples, get out the pen, help us to submit digital recording system related to your stories

regardless even if your fitting went awry during the middle section. Once you'd got to the end of the track you could fix the mistake by either manually cutting and moving the notes into place, or by copying a couple of bars of good bass from another part of the track and replacing the bad parts with this. You could even take that little slapped bass fill from the start and have a go at dropping it into other parts of the track as a little fill, or maybe loop it to make an interesting, modifiable eight section.

A huge weight is lifted from your shoulders when you know you have the ability to edit your recordings in this way. Now, instead of getting your head down and gritting your teeth until the end of the track to make sure no mistakes are made, you are free to relax and enjoy the whole thing, assured by the knowledge that any imperfections or mistakes can be fixed. It also offers plenty of opportunity to drop the more mundane parts of the job of

recording and let the creative juices flow.

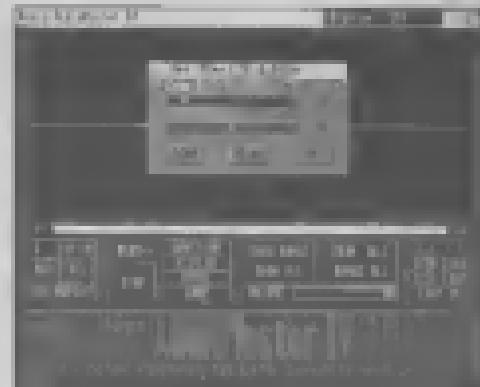
MIDI Sampler

Maybe you already have another computer or a dedicated hardware sequencer controlling all of the instruments in your studio. The Amiga's sampling and sample replay facilities can still be exploited by getting it to act as a sample replay unit triggered via MIDI from your sequencer or a MIDI keyboard. This will allow you to play up to four samples at once. Technosound's Turbo II Pro from New Dimensions has options designed to do this. There are a number of situations in which this could be used. If your equipment is being separated from a computer, whether it's an Amiga, Mac or PC, an additional Amiga acting as a MIDI triggered sampler will fit into the system with no problems.

It is worth remembering that even the oldest A1000 has exactly the same sound chip as the A1200. You can pick up a second hand A500 for next to nothing, and if you need an extra sample replay unit this is well worth considering. Late musicians could fire samples from an Amiga using a MIDI keyboard, although the prospect of lugging around an Amiga and monitor for the sake of four samples isn't appeal in many respects now that there are quite a few portable MIDI samplers available.

Features effects

In any musical setup, effects processors come in very handy. Effects processing is used extensively in professional recording studios, to the extent that many musicians and producers feel completely naked without them! While many sounds and instruments are fine just as they are, certain effects can drastically improve a variety of sounds. Reverb (short for reverberation) is often overused but gives a sound an audience as if it was being heard in a



large hall or a church. This is especially useful for vocals and soloists. Effects can sound rather harsh without any effects and adding reverb to them can also help mask minor imperfections.

The Amiga can be made to act as an effects processor using one of a number of samplers. Almost all sampler cartridges come with software that has some form of reverb effect option.

The best of the bunch is Hobbs' *Amix*. *Amix* is a combination of a 12-bit sampler cartridge that plugs into an A1000 or A1200 via the PCMCIA slot and some clever software. The *Amix* software has unique reverb sound processing routines, which as well as offering 12-bit input and output with up to 128-bit bandwidth for excellent sound quality, also is the only software available that offers convolution effects. For example you could process a sound with both reverb and phaser effects simultaneously. The range of effects processor is very

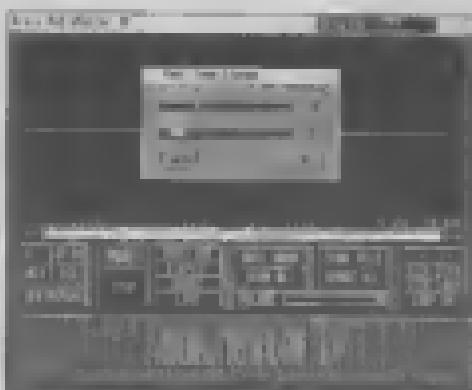


Effects menu for the Amiga 1000 expansion card. It's very similar to the main menu, with a few more options and suboptions.

impressive and the results are a lot clearer than you would get from 8-bit alternatives.

AmicoMaster IV (see cover review on the November 1993 issue of *CU Amiga Magazine*) has a few good reverb echo and reverb effects, although like all of the 8-bit samplers, the reverb routines are rather crude extensions of the echo routines. There's more to creating a convincing reverb sound than repeating the signal at lower volumes. For this one, PicSoft's AMAG is the only package that has a realistic reverb effect. Technosound Turbo II Pro has a mass of reverb effects, many of which are noisy but fun all the same, such as the pitch-changers and spectrograms. Hobo's Megalibound has some reverb effects built in and also comes with a bonus program called The Reverb Effects Generator (REEG) for short.

TRSC was included on a cover disk with the October 1994 issue of *CU Amiga Magazine*. It's



Same menu, different effects. It's interesting to see how much the Amiga can do.

cool program that is completely devoted to realistic sound processing. Along with reverb and delay processes, it also has a good range of echoes and delays. One of the cool ones is the longest delay which feeds the echo back onto itself to create almost endless trailing or ringing sounds from a single bass pulse for the abstract audience.

Unwanted noise is always a problem when using anything less than 24 bits for effects processing, especially when subtle reverb and echo effects are employed.

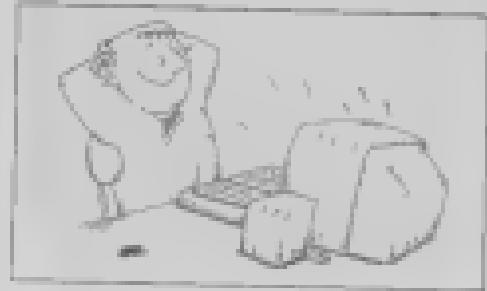
The Amiga's low pass filter can be switched on to clean up these effects.

If you find you don't have the right process available from your reverb effects menu, one way around the problem is to sample the sound you want to process and effect it using the non-reverb effects. Most sample editors will have alternative effects available for processing samples. For exam-

ple, you are unlikely to find a time stretch effect in any real-time processor, but AcidBlaster DV and Acid Illusion can both time stretch a pre-recorded sample. You can then incorporate your altered sample into your sequence during playback.

If you want to use an Arranger for real-time effects processing whilst sequencing from an Arranger you will need to Arranges. You won't be able to do both from the same Arranger because the audio channels will be in use by one application, locking out the other.

Perfect sampling



Anyone who is serious about making music with their Amiga needs to know a thing or two about sampling. The Amiga's four channels of 8-bit audio are capable of producing stunning sounds that are far superior to the results many would think are possible from 8 bits. The reason that 8-bit audio has got itself a bad name are twofold. First of all it is easier for the beginner to make bad sounding 8-bit samples than good ones, and secondly the rest of the world has been working with 16-bit sound as a measure base for years now.

The all-harm attitude often expressed towards 8-bit sound systems is not undeserved. Frankly it takes quite a bit of experimentation, practice and experience to squeeze 'professional quality' samples from an Amiga, which is why we've devoted a major section of this book to exactly that cause. If the man, called for your Amiga makes an attempt to be clever, modulus collections, patterns and so on, then nothing

more sampling techniques are available. It will also surely help. On the other hand, you represent a company trained for commercial release. Whether and to a cassette or vinyl, the final sound quality and production is crucial.

It pays to set your sights high when it comes to production quality. A clean, bright, punchy recording made from your Alesis onto a DAT tape, can be used as a master recording for pressing a single, EP or an album. This gives you the power to either release your own music or present a record company with a ready made master tape. Let's say for example you had made a killer record, but the production was fuzzy and lacked definition. You tried it to a record company and they're uninterested - then ask you for the master tape and you have to tell them that the tape you sent them was the best you had. If they were still interested in releasing the record, there would be no option but to record the song in a professional studio, which would almost certainly not be equipped with an Alesis and your specific tracker software so you would have to try to reconstitute the record on an amateur system, reprogram all these fiddly bits and maybe lose the whole value of the track. You might also have the bill for the studio time deducted from your royalties, or advance from the record sales. Gah! So to get your production right from the start, begining with perfect samples.

Sampling rates

The biggest factor governing the overall quality of a sample is the rate at which it was recorded. This is known as the sample rate. An explanation of how sampling works can be found in Chapter One.

Depending on what sound you are sampling and how you intend to use it, the best sample rate will vary. If you are low on Chip RAM or you have



Based on a 16 bit digital recording, a 16 bit digital sample at 44.1KHz will contain 16 bits of information per sample.

other memory limitations you will want to use the maximum sample rate you can get away with. This will use as little memory as possible. If you use too low a sample rate the upper frequencies of the original sound will not be captured. Instead you will get a poor substitute which sounds hard and scratchy. However, let's assume you have a generous amount of Chip RAM to play with (128 is standard on the A1200). Unless you are using lots of very long samples in series of two or four bar loops for example, this should give you enough sample memory for most projects.

If you do not intend to re-use your samples then by all means use the highest rate you can. However, just because you have a lot of memory space, it's not necessarily best to sample at the highest possible rates all the time. Trackers work by re-playing sampled notes at different speeds, which changes their pitch. Generally the higher sample rate a

samples to replay at 28kHz – this is the rate at which you play the note A4 (note A on the third octave). If you sampled a single note at an intermediate 28.2kHz, when you loaded it into a tracker you would not be able to play that note any higher than the original pitch, so in the case of sampling single instrument notes you need to sample at a lower rate so that it can then be pitched both up and down from within the tracker.

C.2 is the note in a tracker that corresponds to a sample rate of 14kHz. This is a good rate for sampling most instrument notes. If you are using a tracker and you want the notes on the display to match the notes that are playing, make sure that the notes you sample match their appropriate sample rate. If you can, tell the pitch of the original note you are sampling, compare it to the following chart to pick a suitable sample rate.

Suggested sample rates

The following list suggests minimum sample rates for a variety of sounds. The trade-off between sam-

ple rate, sound quality and memory use is open to variation, depending on your requirements, but these figures have been reached on the basis that you do not have location memory but high fidelity is a priority. As they do not need to be released, the rates suggested for drum loops and one-shot sounds like drum samples can be increased to the highest your memory will allow with an according decrease in reproduction quality. The scientific method for finding the minimum sample rate is to calculate the Nyquist rate. This is twice the frequency of the highest pitch in your original sound. The easy method is to use the list.

Sound	Suggested rate (kHz)
Tuned instruments	16,712
Voices	22,272
Single drum sounds	22,272
Drum loops	22,272
Triangles and extreme high pitches	22,168
Base notes	16,712

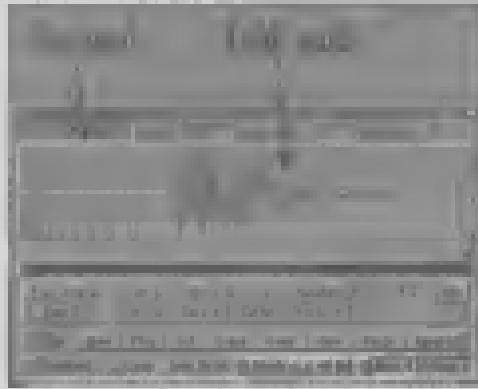
Recording waveforms

One of the biggest advantages the Amiga has over dedicated MIDI samplers is the fact that you can actually see the sampled waves on screen, large as life. If you've ever moved from vocal sample editing to using a MIDI sampler that has nothing but a numeric readout, you'll know what an enormous aid that is to sample editing. The key to getting the most from graphical sample editing is learning to read sound waves.

When you examine a sound wave for the first time it normally looks just like a random series of spikes and lumps. However, if you use the zoom controls on your sampler editor to play and examine specific parts of the wave you should start to see a

Note	Period (s)	Note	Period (s)	Note	Period (s)
C.1	0.00	C.1	0.0	C.1	0.000
C4#	0.01	C4#	0.01	C4#	0.000
C.1	0.02	C.1	0.02	C.1	0.000
C4#	0.03	C4#	0.03	C4#	0.000
C.1	0.04	C.1	0.04	C.1	0.000
C4#	0.05	C4#	0.05	C4#	0.000
C.1	0.06	C.1	0.06	C.1	0.000
C4#	0.07	C4#	0.07	C4#	0.000
C.1	0.08	C.1	0.08	C.1	0.000
C4#	0.09	C4#	0.09	C4#	0.000
C.1	0.10	C.1	0.10	C.1	0.000
C4#	0.11	C4#	0.11	C4#	0.000
C.1	0.12	C.1	0.12	C.1	0.000
C4#	0.13	C4#	0.13	C4#	0.000
C.1	0.14	C.1	0.14	C.1	0.000
C4#	0.15	C4#	0.15	C4#	0.000
C.1	0.16	C.1	0.16	C.1	0.000
C4#	0.17	C4#	0.17	C4#	0.000
C.1	0.18	C.1	0.18	C.1	0.000
C4#	0.19	C4#	0.19	C4#	0.000
C.1	0.20	C.1	0.20	C.1	0.000

You will notice that the sample period is a major part of the note length in a 1/16th note pattern. In full musical 1/16th note length, the note duration approaches 16 notes. There are notes in this chart that are.



new patterns emerging. Loud parts of the sound coincide with parts of the sample wave that vibrate the very top of the wave display to the bottom while quiet parts are signified by a much smaller wave that oscillates along the center line. Also, parts of the sample that contain high frequencies, such as cymbals and drums, will be displayed by a lot of very tightly packed spikes in the wave display. At the opposite end, low frequencies are shown on the wave form as long smooth curves. A sample of a bass drum and a cymbal would have the long wave of the bass drum broken up by the spikes of the cymbal. A complex sound, such as the human voice, will appear as a succession and combination of various frequencies.

Take a look at the sample waveform and then compare them to some of your own sounds in your sample editor.

Once you know what sounds look like you'll be

able to trim down a sample to exactly the part you want in a matter of seconds. For example, if you are sampling a one-beat drum loop from a sample CD, you'll often find that the first note of the following bar has been included on the CD, so if you were to loop the whole thing, it would ping and sound out of time. When you know what you're looking for, chopping off the last beat can be done without even listening back to the sample.

Removing noise

Noise can also be seen on a sample wave to great effect. Noise in this case does not mean the same thing as sound. Noise is any unwanted frequencies in the recording, which usually takes the form of high frequency hiss or low frequency boom.

If you have a sample which is lived at the start then has a short silent gap and then has another



With a bit of practice you will be able to see unwanted portions of your samples and to remove them. Here the highlighted section is removed from the sample waveform.

load part to rhythm guitar loop for example) you may find that bass or bass noise creeps into the bit that's supposed to be silent. This will look like small but rapid oscillations travelling along the middle of the waveform display. The same noise is probably present on the whole of the sample but will not be so obvious when the volume of the original sound is substantially greater than the volume of the noise.

So removing this noise from the silent part is required. There are a number of ways you can do this. You could take the relevant section, to remove high frequencies in the case of bass, or low frequencies in the case of bass. This is the best route if there are any traces of the original sound in this quiet part, as they should remain fairly unaffected by that filtering. If all you want is complete silence in the gaps, the easiest thing to do is highlight the noisy area and reduce its volume to zero. Even the most

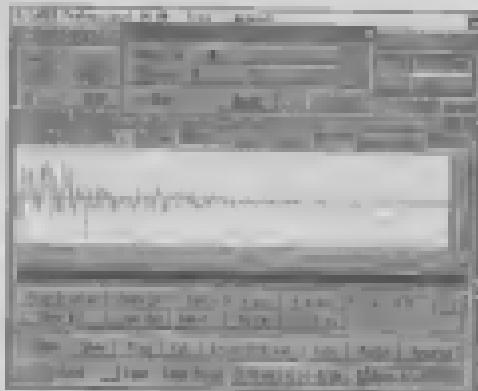


base sample editor will have this ability. A tip for users of CVP's D9500 sampler is to turn on the anti-walk filter (see the Amiga low pass filter) just before sampling base sounds. This eliminates almost all high frequency noise from the sampling signals.

Avoiding noise

Of course, the best way to avoid noise samples is to remove as many possible sources of interference as possible before you sample. Noise can leak into the signal from a number of places. The main safeguard is to make sure you have a good strong signal entering into the sampler by using the transfer function of your sampler to check that the sample wave just very occasionally touches the top and bottom of the waveform display without flattening out. This will give you an optimum signal to noise ratio where the volume of the signal is significantly high, other than that of the noise.

Ironically most of the noise you will encounter will generally come from the computer and music systems themselves. Four culprits for noise pollution are TV's, monitors and internal hard drives. The best way to test for noise generated by your system is to sample some silence with nothing connected to your sampler's input. If you play back the sample and there's noise you can start a process of elimination, repeating the process with various bits of the system disconnected until you find the culprit. If you really care about clean samples, and you have, for example, an AEG with an internal hard drive, it's worth disconnecting it and booting from floppy disk to see if there's a significant improvement in sound quality. By the way, these devices will mainly affect the quality during sampling, but it's also worth turning off non-essential devices if you are recording a master tape from the *Amiga*.



Boosting frequencies

While filters are useful things for reducing noise, boost options can be just as useful for enhancing the frequencies that you want to focus on. For example, boosting the high frequency content of certain drum samples can make them sound much brighter and cut through the mix. Likewise, the lower end of the spectrum can be boosted to give more oomph. Creative use of filtering and boosting specific frequency bands is one way to inject new mood and flavour into drum loops and other sounds.

Invisible looping

In some cases you'll want to add loops to your mix. For example, a sustained environmental sound such as a train could be looped and held for as long



Invisible loops can be created by looping a waveform by setting markers at the start and end of the loop, and then looping the sample between them.

as required. A looped sample will have two markers: one for the start and another for the end. The long start marker doesn't have to be at the start of the sample so you can play a sample so that the first bit plays just once. Then the parts between the loop markers play indefinitely. The tricky part here is finding the right loop points so that the sample continues without either jumping or clicking.

First you need to get the hang of looping with a click. A click occurs when the waveform jumps very suddenly from one position to another. Open the bottom of the wave to the top for example. Attacking this is fairly simple so long as you make sure that the start and end markers are both positioned at places where the waveform crosses the middle line of the display. An alternative method is to use two points where the wave reaches either the top or the bottom of the display.

Creating a natural sounding loop requires a little

more techniques. For this you need to use both your ears and your eyes. What you need to find is a place on the sampled wave where the sound *edges* continue with almost exactly the same sound for a while, or where it wobbles back to something like the first part of the sound. You need to find a smooth join. It's a bit like putting up wallpaper when you have to line up the pattern of the new roll with the one you've just put up. The best way to do this is by listening and examining the sound wave, looking for sections that look similar. When you think you've found a suitable pair of loop points, set up your markers and try them out. If you're unsuccessful you'll have to keep on trying!

There is a way of cloning. You can copy a large section of the body of the sample (the part you want to loop), paste it onto the end of the sample, then reverse just that section you've pasted on. The result should be a 'butterfly' effect, in which the sample plays and then the second section is played again in reverse. If you've been careful with your cutting the join should be free from any glitches or clicks. You can then loop to interested part of the sample.

Sound Effects



Sound effects crop up everywhere: decision games, presentations, soundtracks, and even the Windows likes to beep and bleep until you tell it to stop. If you're interested in developing any of the above, then the ability to create your own sound effects is vital. Even if all you want to do is add a sound sample to your boot-up sequence.

The techniques and processes involved in creating sound effects can be different for each particular application, but we'll start with a look at making standard MP3 sample sound effects which can be used in almost all of these cases.

Sampling CDs

Obviously if you're using sample sounds, the question and answer will be: get results is to find the sound you want and sample it. This is referred the best option in many cases, especially when individuality is not a high priority. If you decide to take

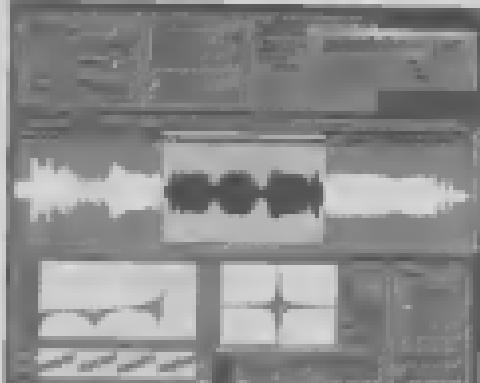


Figure 1: A digital audio workstation. There is no sampling function in most digital audio applications but this has to be done manually.

thus approach you then need to find the sounds you want to sample. Video tapes are a common source but there are associated problems. First of all there's sound quality - VHS is not known for its high fidelity audio and anything but a first generation copy will usually be extremely noisy. The other problem regards copyright. Although this won't be important if all you are doing is making a sample to your Workstation, you will still be in breach of copyright law and hence breaking the law if you record a sample from a movie. This will have to take into consideration if you are working on sound effects for a game or anything else that is to be made available to the public. Then again, this would not really apply if you took just a split second of generic sounds rather than a whole lot of recognisable speech (although technically it would still be an illegal breach of copyright).

The best source for sound effects samples is a

dedicated sample CD. Sample CDs come in both audio and CD-ROM formats and are usually filled with thousands of different sounds presented in a samples friendly format. These CDs can be separated into two types. The first will have had no content added for copyright infringement or will include totally originally generated sounds, or either way you are free to use the sounds in your own production without fear of lawsuits from the sounds original creators. The other type is more common and will contain sounds from a variety of sources. These may contain samples from copyrighted recordings that have been reproduced by the sample CD manufacturer without the consent of the copyright owner. In this case the producer of the sample CD is breaking a number of copyrights although in most cases the CD producer gets away with it although not always. You may need to gain permission for the use of sounds from such a CD from the original copyright holder. On other occasions the producer of the CD may have cleared samples for release on the CD but this does not necessarily mean that they are free to use them at will.

Copyright law is a rather grey area when it comes to sampling parts of existing works for use in a new piece. The way to get away with it is to alter the original sound to such a degree or use it in such a way that it is no longer recognisable in its new surroundings. As far as sound effects go, copyright problems are only likely to arise with the use of phrases of speech or particularly unique and well known sounds. Be safe, always check.

Sample Rippers

There are other sound sources closer to home. Many games and demos will have their sound data stored in non-standard formats making it impossible

be to load their samples and soundtracks into one revisionable audio program. However, when you extract your Amiga, much of the memory will remain unchanged and the sounds could still be resident in RAM. Sample editors and module editors allow you to scan the memory of your computer for any samples or music modules that may be present. In the case of module editors the utility will tell you if it has found anything it recognises and ask if you want to save it out to disk. You can then load it into the relevant software and "borrow" its samples for use in your own work. Sample editors work along the same principle, but this time it's up to you to decide when you've found a sample. Most sample editors have a RAM Scan option which allows you to replay the contents of the RAM as if it were a sample. Most of this will sound awful but when you locate a proper sound you can highlight it and save it out to disk.

Any samples you obtain this way will be subject to all the normal copyright limitations, so it's best to alter them with some of the following techniques to avoid any trouble and make them 'original'.

Sample editor tricks

Everyone wants their sound effects to be original - to make the listener sit up and take notice. Once you've got your basic sound editing in your sample editor there are a number of tricks you can use to spice it up and squeeze extra value from it.

Looping effects. This isn't so much about being tight with memory but more to do with giving yourself more scope for extra sounds. For example, if you can reduce a 32K sample to 1K with no loss of detail or sound quality, you've just earned yourself an extra 23K for a brand new sound effect. This kind of memory saving isn't always possible and in many cases you'll find that what seems to be a fairly



Sample editor software allows you to scan your computer's memory for sounds that can then be used as samples in your original music.

long sound is in fact a very short one that's been looped. Listen very carefully for repetitions in your sound effect as these will indicate a looped sound. These repetitions may take place extremely rapidly and could be too fast to hear so try replaying your sample at half speed. Also take a look at the waveform. If it follows a repetitive pattern then you can take just the first instance of the pattern and loop that, chopping off all of the rest of the sample. Sounds that are likely candidates for looping include background noises like the bleeps and hum of the Starship Enterprise or the chug of crackles in the pongish wagon wheels, white noise (such as wind and waves), machinery running over and many others.

Remember that if you are going to use these samples from within a tracker or one of your own programs (C, Assembler, Basic, Pascal etc.) you will be able to alter the volume and pitch of the

loop at that stage, which will greatly increase the search possibilities. Check the Programming and Techniques sections of this chapter for more on this area.

Phasing: Phasing is a side effect of mixing two copies of the same sound together where the patches and speeds of each copy are slightly different. The resulting sound will be similar to the original but will have a "whoooshing" sound that appears to move up or down in pitch. This is very useful for adding movement to a sound, giving the impression that the subject is moving across the scene. As an example, have a go at making your own jet plane fly-by sound effect. You'll need a "white noise" sample that should be a few seconds in length. If you have Kontakt ED you can use the Kontakt Noise replace from the sample editor to create some white noise from scratch. Never save the sample and load it into a sample editor that has a phase function.

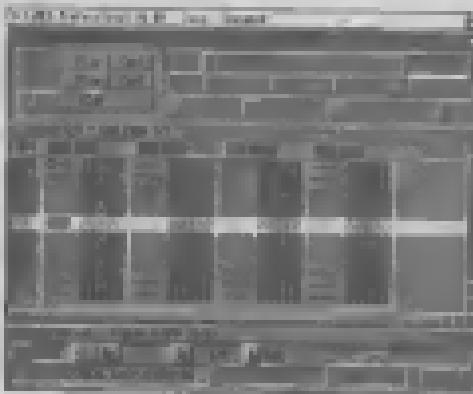
AudioMaster IV (from the November 1993 issue of *CLT Anaga Magazine*) has this in the form of its *mix function* – just adjust the *Other Value* to anything you like. Position the slider on the white noise and you'll get a dramatic swooping sound as if a plane had just flown overhead. Experiment with phasing effects on any other sounds you want to process.

Echoes: Echo effects are very useful if your sound effect sample is cut off suddenly at the end. By adding an echo to it, the abrupt ending is smoothed over with little repeats that gradually fade away. A nice original use of the echo function is to set the echo rate at a very high figure for a very low one, depending on how your sample editor works. The idea is to get the echoes occurring in very quick succession. If the echoes are close enough together the result is a metallic sound rather like the effect used to process the Edgesong effect. This is one of

the easiest ways to diagnose your own voice. And if you have the misfortune of regional accents you can record tough dialogue samples worthy of the most fearsome interlocutor.

Tracker box

Years of you're not wanting certain tracks are very useful for stringing together a series of sound effects. The first control over the samples that tracks are often used for many sound effects applications. For example if you want to use the techniques explained earlier and loop a small sample for use the sound of machinery engines, laser guns or whatever you can re-establish the original sound with reverb effects such as pitch slides, volume slides, vibratos and arpeggios. By combining volume and pitch slides, a tiny sample loop can be transposed, if into a full blown great sound generator.



appearing from the distance, reaching and leap-
ing and shooting off into the distance. It's well
worth investigating this a route if you need to add
sound effects to an animation. Trace the animation
and make notes of which sounds need to be used
and faded at which specific times. You can then
construct your sound effects module accordingly.
The animation and sound effect module could then
be deployed from within a multimedia authoring
package such as Videolope Pro from the March
1996 issue of *CLi: Amiga Magazine*.

Programming tips

For those who like to program their own games and dances, the possibilities are greater still. First of all there's the full control you have over the basic sample/replay parameters (patch, volume and so on). In addition, you have the ability to do much more.

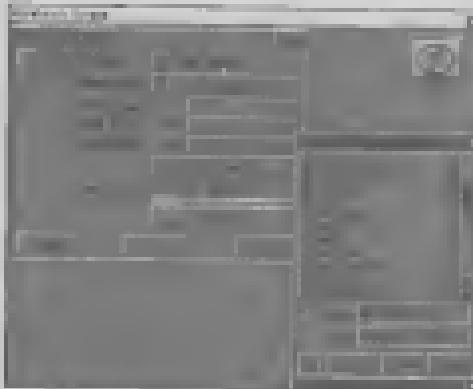
With the program, *Algorithmic Music*, number generators can be programmed to add complex and unpredictable angles to your sample's pitch and volume, and these alone can transform it sound. With some imagination and inventive programming, you can synthesize whole new sounds from what were previously ordinary samples.

Workbench examples

Adding your own samples to your Workbench is very simple. Recent versions of Workbench come with a Preferences screen dedicated to the user as shown. This allows you to select either a sample beep or a specific sample which will be loaded during the boot sequence. For that reason if you don't have a hard drive you'll need to include the sample file on the Workbench boot disk. You can set the replay rate by moving the slider. The *Use File* button



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to find the right reply speed, then click the **Save** button to make the changes permanent. The selected sound will now play whenever you hit an error or make an unsatisfactory selection from within any system compliant application on the Workbench.

5 Sequencers and Trackers



There are two families of music-creation software for the Amiga: trackers and sequencers. While there is some common ground, the two types work in very different ways and are suited to different situations. Before we go into specific detail about any one particular tracker or sequencer we should be sure of exactly what constitutes a tracker and what makes a sequencer.

The tracker

The tracker was originally developed as a way of creating high-quality music for Amiga games and demos. The final product had to be very efficient with memory and processor power but also take full advantage of the Amiga's then revolutionary sound hardware. Soundtracker was the original tracker developed by programmers on the Amiga demo scene. Third parties took it upon themselves to continue the development of Soundtracker by

dition, the name is unique to the program and re-distributing it. Some of these extended versions went out under the name SoundTracker but many altered the first part of the name to reflect the changes in the program. Before long, there was the choice of SoundTracker, NoiseTracker, StarTrekker, Freetracker, ChaosTracker, Faketracker, as opposed to the original and by-then way out on the air. Then there were the slightly more unusual tracks on the theme, such as MovieSister (MIDI), Game Music Creator (GMC), Oktalyzer, TPMX, and more.

More than anything else, the one thing that characterises a tracker is its, er, vertically scrolling alpha-numeric display. This is normally the biggest stumbling block for anyone approaching a tracker for the first time. The bank of numbers and letters have no relation to traditional music notation and there are no descriptive headings or guides to be seen. Why? Because the tracker was never supposed to be a tool for the traditional musician, and any compromises it made to address that would reduce its use as a compact musical tool for games and



games. Power over Paula is what makes the tracker such an attractive option. Most trackers have a sequencer editor built-in, so you can grab new sounds half-way through making a song and edit them to your requirements. The events control over everything from precise volume levels, pitch bends, arpeggios and slides gives ample room for sonic experimentation. Quantisation is very rigid unless the feature is turned off, which makes creating patterns in 'trip time' a very quick and easy process, although the latter trackers will counter this with options to achieve notes that are just off-beat, or triplets if required.

The beauty of a standard that's been around on the Amiga since the start is that the MOD files that are created by trackers are called modules; supported by virtually every piece of creative software that you'll ever find for the Amiga. MOD files can also be loaded and played by a large number of PC and Apple Mac programs. Over the last few years programmers have been developing alternatives to the MOD format with extra features such as 32 or more voices, 16-bit output, MIDI sequencing and more, so in fact there are quite a few 'standards' of the MOD format.

The main contenders

There are many variations on the tracker theme for you to choose from. Here are some of the leading contenders for your attention.

ProTracker

ProTracker is the choice of the majority of musicians who write primarily for games and demos. An accomplished programmer will be able to slip a ProTracker module into a game or demo without breaking an eyelid. ProTracker conforms to the tracker tradition of a non-standard interface, there are no

pull down menus and all the buttons are arranged in a rather poky collection of banks, some of which change to reveal alternative banks if you click on the right one. Once you know your way around the interface there shouldn't be too many problems, but the instructions the program makes about your Amiga, its disk drives and its filing system can be frustrating.

You won't find any MIDI support or many eight voice features but there is a sample editor. ProTracker is and I would say for a couple of reasons. First and foremost is familiarity - I have been using it for the past five to ten years, changing to an alternative system isn't an attractive option. The other reason is simply because it is widely used, which in turn means that there is plenty of support from the rest of the Amiga scene. However if you're new to it all, you may be better off trying out one of the more general sequencing alternatives.

ProTracker is public domain, the current version is 3.01, and it's available from all good public domain software suppliers.

OrchMED

Any regular reader of CU Amiga Magazine will know all about RSP Software's ChordMED since it keeps it listed. At the time of going to press the latest release is Version 6, also this is soon to be superseded by the next update, OrchMED Soundtracks V7. Originally based on the tracker theme ChordMED adds a number of power features such as comprehensive MIDI sequencing alongside the samples, synthesis sound generation. A standard Workbench 2 interface with menus and windows and full ProTracker compatibility. Soundtracks will offer more, including up to 64 channels of Amiga samples that can be output through the five 16 bit sound card with full stereo panning controls and

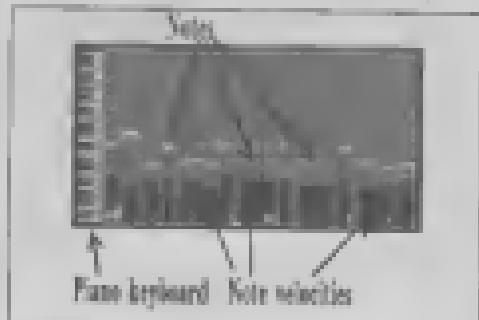
volume solo effects. If you want to combine MIDI sequencing and powerful Amiga sample control, ChordMED is definitely the way to go.

Musicleve Editor

One of the most original twists in the tracker theme has come from Musicleve Editor (VHSine for short). The main attraction here is that you can create instruments by loading in samples and then assigning a number of synthesis values to them. For example, you could load a sustained string sample and give it some resonance, which would change a fairly ordinary sample into a more interesting sound with its own movement and character. At the moment VHSine is still in its early stages of development but it's worth checking out. With more support from users it could grow into a force to be reckoned with. Musicleve Editor is shareware, available from any good public domain source.

Sequencers

In contrast to a tracker the main aim of a sequencer is to present the musician with a quick



and every way to record, edit and replay music from MIDI instruments (although Amiga samples are usually supported to some degree).

Sequencers have no place for scrolling lots of numbers and letters. They prefer to display their musical data in the form of a piano-roll display. This takes its name from the rolls of paper used to store musical scores for self-playing pianos. A piano-roll display is read from left to right using a horizontally scrolling reader. Along the left edge of the display is a representation of a piano keyboard stood on its end, so the lowest note is at the bottom and the highest note at the top. The piano is used to indicate the pitch of the notes, rather like the numbers running up the side of a bar graph indicate the values of the bars. So each note is given a vertical position according to its pitch, while its duration is represented by the length of the note's bar. In addition to this, each note may also have a vertical bar which indicates its volume or velocity.

A typical use of a sequencer would involve a setup where there was a MIDI keyboard plugged into the Amiga, with the keyboard acting as both an input and output device. The sequencer would be set to record on track one, while the Amiga played a beatline along to the tick of a metronome. When recording had finished, any badly timed notes could be quantised and corrected with a quick mouse click or menu selection, then the next part of the music could be recorded onto track two (perhaps a chord sequence). The sequencer only records MIDI data, which contains information about which notes were played when and for how long. When everything has been recorded it can all be played back through the keyboard via remote control from the computer sequencer.

Unfortunately while there are quite a few sequencers available for the Amiga, development

has stalled on all fronts. The advantage of this is that you can pick up some very powerful software for little cash if you shop-around.

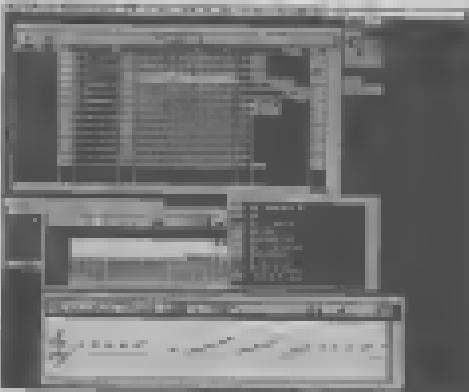
The competition

How are the main players in the field?

Bars and Pipes Professional

Bars and Pipes Professional is without a doubt the king of Amiga sequencing, offering a range of features, plug-ins and overall power that knocks the competition for six. Bars and Pipes likes a challenge. It is known as if it's saying "Go on, give me your best shot, your biggest MIDI set-up, your most demanding music project ever" and it still manages to take it all in its stride.

Much of its success lies in the system of plug-ins and controllers. These can be used to add creative



The big brother of all Amiga sequencers Bars and Pipes Professional is unique and powerful, so here's the sequencer that can do it.

newer effects to your sequence or to integrate additional hardware seamlessly into the sequencing system. For example, there are specific plug-ins for direct links with Triple Play, Play 64-way MIDI interface, Sonoma ADSR16 and Tascam 16-bit hard disk resident and plenty more. Bass and Popcorn is at its best when run on a well stocked rig for Amiga and a large feature free interface to take advantage of its excellent display features that are based around a colourful set of icons and windows.

Music X 2.0

Music X 2.0 is a much on the rest of the Amiga sequencers by simply being there at the beginning. Although it can't hope to compete with the bag of tricks that is Bass and Popcorn, it has plenty of its own to tempt the Amiga userbase. The fact that it doesn't have a massive array of options is a positive



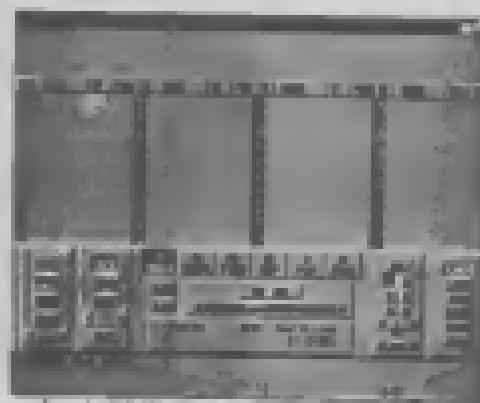
Music X 2.0 is a much on the rest of the Amiga sequencers by simply being there at the beginning.

point for those working on lower spec'd machines and also those who like to work in a more simple packaged environment.

There are some nice recording and structuring methods on offer in Music X which allow you to record parts and use them in subsequences in the overall sequence. Many find that the features for building Amiga samples are the best of any sequencer, so if your basic needs to be primarily MIDI-based but you still want good control over Amiga samples this is probably your best option.

KCS 2.0

Dr T's KCS (Keyboard Controller System) went over well back a few years ago but you may well find this very powerful sequencer gathering dust on a retailer's shelf somewhere. The last release was a combination of KCS and Bass sequencer! Tiger



For any basic sequencer, KCS is a good choice. It's not the most complex system but it's got a great

Cub (the display system) and *Autodesk* (an automated MIDI control tool).

This system could not be described as user-friendly but it does offer a good deal of control over a large MIDI setup. The most interesting part is the automated mixing desk. This allows you to record volume changes for as fast changes of any MIDI controller via a mouse-controlled virtual mixing desk. If your real mixing desk is equipped with faders, you can even control those from the program's *Autodesk* section.

Amiga audio hardware add-ons



8-bit samplers

Most Amiga 8-bit Amiga samplers follow very similar lines. They consist of a stereo sampling cartridge which plugs into the Amiga's parallel port and some software which reads the data from the cartridge and allows you to edit and save the samples. There's very little to choose between the hardware cartridges, so most buying decisions are made on the strengths or features of the accompanying software. Most sampler cartridges and software can be interchanged with different boards. For example you can use the Megadiscord cartridge directly with the built-in sampling software in Octetrix PDI. All of these use the Amiga's own hardware to replay the sampled sounds. The samples created with these packages can be used in any Amiga music program.

Aura 8

HiSoft £94.95

When HiSoft developed their original Aura sampler four years ago on 16-bit samplers they came up with some impressive editing software. Now the software is available for 8-bit users in the form of Aura 8. The hardware is a small cartridge that plugs into the Amiga's parallel port while the software is basically the same as for the Aura 16. For the price this looks like an excellent option for anyone who wants a bit more than a simple sample grabber.

Megalosound

HiSoft £24.95

Megalosound is a neat little sampler package that comes with some good software. The hardware is a stereo cartridge that plugs into the parallel port as usual but also has input volume control. This is particularly useful as it means you can sample direct from the outputs of a CD player bypassing the potentially noisy amplification stage. You could do this with any sampler but the volume control ensures that if you are sampling from a very loud CD you can avoid distortion.

The Megalosound software has some quirks in its interface but for some reason this seems to be par for the course when you're dealing with Amiga sampling software. There are also some useful effects available for processing samples, especially if you like distorting sounds to creating new ones. Excellent value.

Technoscound Turbo 2 Pro

Emerald Creative £26.95

Technoscound has been around for years. Its latest incarnation seems to sporting a large number of just for fun/realistic effects. Unfortunately the sound quality of many of these means that they really are just for fun. There is a built in tracker which makes it

relatively easy to try out your samples in a musical context, although it is very basic and not compatible with the MOD file format. On the subject of the mod file format, Technoscound is the victim of a bad bunch. Nevertheless, it has its fans.

OS8 8 Plus

Power Computing £19.95

Originally developed by the highly respected GVP in the USA, OS8 8 Plus has a clear plastic cartridge with its internal components on show. The highlights of the software include a unique pre-sampling high frequency filter. This is completely separate from the Amiga's own low-pass filter, and when enabled an before sampling it removes bass from your incoming sound before it is sampled. This is very useful for sampling bass sounds, and the resulting samples are surprisingly pure. A tracker is included in the program and although the software is low on effects processes it's one of the most user friendly editors.

16-bit Samplers

Things are rather different when it comes to 16-bit samplers. As the Amiga has no 16-bit recording or replaying facilities, each sampler takes its own approach, interfacing in different ways and using specifically designed hardware and software. This means that in order to use the resulting samples in other software, the software will need to have special support for that particular sampler and its files.

Aura 16

HiSoft £79.95

The cheapest route into 16-bit Amiga audio is Aura 16, although it does it actually offer 16-bit sampling or replay - instead it samples at 12-bit resolution and converts this up to 16 bits during the editing and processing stage and then back down to 12 bits for

output. Auru connects to an A1200 or Addura via the PCMCIA slot. There are stereo inputs and outputs on the cartridge, with stereo pass-through inputs for combining your standard Amiga audio with the Auru output.

One of the best things about Auru 16 is the selection of high quality reverb effects that are available from the software. These are especially remarkable because they allow you to combine more than one preset at a time, such as phaser and echo for example. Auru also has excellent sample processing features. Third party support is available in a limited form from OctaMED 5.24 upwards, in which case you can substitute one Amiga channel for a mono Auru 16-bit channel.

Toccata

White Knight Technology £2999

Toccata is a Zone card for all 'big box' Amigas. It has been designed by MicroDesign specifically for use as a hard disk audio recorder with their V-Lab Motion card in a digital video editing set-up. In which case it handles the soundtrack while the V-Lab Motion deals with the pictures. Samples can be recorded to and played back from both RAM and hard disk. Toccata offers stereo sampling at rates of up to 16MHz or 16-bit. Unlike most samplers, the sampling and replay rates must be chosen from one of a number of presets. These include 44.1kHz (perfect for mastering to CD) and lower rates such as 32kHz (the rate used by Amiga 500/16 level MIDI sampler). This means that normally it cannot be used to sample a single note and then use this to replay a melody since there are not enough presets to cover all the notes that would be needed.

However, when used in conjunction with OctaMED 5.24, samples can be built from both 8 and 16-bit samples and fed to the 16MHz 16-

bit output of the Toccata, which gives superior sound quality to using the Amiga's standard 8-bit output. In this case each track can have its own stereo pan position and echo effects can also be added in real-time.

Samphoria is the software supplied with the Toccata, featuring a good selection of basic editing tools and a few effects. Toccata samples can also be string together into a single track sequence using Samphoria's Playlist feature. Bars and Pipes also allows Toccata replay features. If neither SoundBender nor Bars and Pipes is your sequence of choice, Toccata will be of most interest as a hard disk digital recorder. For this job it could really do with some more advanced software to match professional systems on other platforms.

Samphoria ADR16

£180-£220 (second hand)

The first ever 16-bit Amiga sampler was the Samphoria ADR16. Like the Toccata, it's designed as a hard disk based digital mastering system, but unlike the Toccata the ADR16 stores two stereo channels instead of just one. Samphoria ADR16 is also a Zone card, and once again (like the Toccata) it too can be paired with a digital video card, specifically the OME Personal Animation Recorder (PAR). While the ADR16 does have support from Bars and Pipes, it is not of much use to anyone who wants a 16-bit sampler for use as a musical instrument.

When the ADR is really scores in as a hard disk recording, mastering and audio editing system. The latest version of its companion software gives precise control over your recordings and has a very user-friendly play list sequencer that works on a drag and drop principal. For hard disk mastering and digital audio work this is definitely the pick of the crop.

Miscellaneous hardware

There are also some other interesting bits and pieces available to enhance and manipulate your Amiga's sound. The standard line level stereo analogue out put is compatible with all audio recording equipment, but here are a couple of tailor made units that are especially geared towards use with the Amiga.

Sound Enhancer

Omega Projects ESR 95

One of the problems with 8 bit sampling is that it can have the effect of producing slightly dull sounding samples. The Sound Enhancer is a fairly poor bit of kit that is very effective and also very simple. It's a small box that sits between the Amiga's audio-in port and your amplifier or mixer.

It works by boosting certain present bass and treble frequencies. The level of the bass boost cannot be changed but the treble control located on the unit can be used to cut or boost the treble frequencies. This is not a mistake written by any means, and if you are using samples played at live takes then the treble boost will only emphasize the distortion. However, if you are working from good samples the effect can be amazing, giving your top end more sparkle and adding plenty of beef to the bottom end. This is especially useful if you are producing house tracks or jungle, in which case the definition of the bass and treble can be of paramount importance.

At the very least, the Sound Enhancer is a neat replacement for a full multi-band graphic equalizer. Once you've used one for a day you won't want to be without it!

Little Gern Micro Mixer

Emerald Creation (086) 995

Another little problem with the Amiga's output is that when four channels of mono samples are played there is no way of adjusting the stereo pan position

of each (although SoundOuts and other multi-channel trackers now offer panning). Two channels are fed through the left output and two through the right. The Little Gern Micro Mixer is a two channel mixer that offers independent pan controls for each channel, allowing you to mix both sides to the centre or anywhere else in the stereo field.

On top of this, there are independent volume, bass and treble controls for each channel. The added in EQ and pan controls goes a long way to curing one of the biggest problems you'll face when making a master recording of an Amiga sample-based song. It will also prove useful in small MIDI set-ups where there is a shortage of controls or channels on the main mixer.

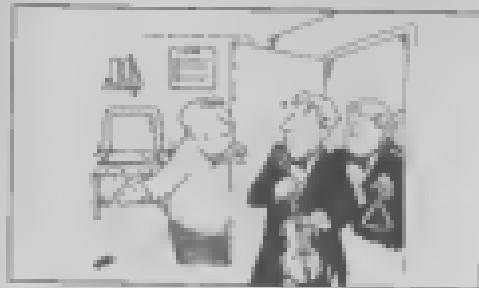
One Stop Music Shop

(150-2200) second hand

A few years ago Blue Ribbon released this unique Zorro sound card. Unlike the Zorro sampler cards covered earlier in this chapter the One Stop Music Shop is really the internal workings of a Pro-Mix multi-sound synthesizer. Conforming to the General MIDI standard it has all the sounds that are specified by GM along with some of its own. Using samples of real instruments as a base for its sounds, it enhances these with an own synthesizer processor.

Anybody looking for a real multi-instrument sound module with a variety of synthesised, orchestral and drum sounds will be satisfied. The One Stop Music Shop can be used with any sequencer running on the same Amiga for externally with the use of its MIDI loop-back mode. This allows your sequencer to talk to the card through a MIDI interface and a MIDI cable, so there is no need for specific support from your chosen sequencer.

MIDI and recording extras



Once you venture outside the realms of the basic Amiga tracker and into the world of MIDI and multi-track recording your options are opened right up. An almost limitless number of MIDI sound modules can be connected to and controlled from your Amiga. Once you start adding MIDI instruments you'll need to get yourself a mixer to combine the different sound sources and while you're making your musical shopping list, how about adding a few effects modules too? The best thing about building your own MIDI system is that you can tailor it to your specific requirements, adding only what you really need (and what you can afford). All of the gear covered in this chapter can also be used with MIDI sequencing systems other than the current range of Amiga computers, so even if you eventually upgrade or replace your current computer, your external equipment will not be redundant.

MIDI

MIDI (Musical Instrument Digital Interface) is the communication standard used by electronic musical instruments. The Arranger has never been factory fitted with MIDI ports but this is not a problem as a MIDI interface is a cheap addition to any system starting from around £10. A MIDI interface plugs into your Arranger's serial port and adds DIN sockets for MIDI In and MIDI Out connections.

The better interfaces will have more than one MIDI Out connection. Even if you only have one port of MIDI equipment when you buy the interface it's still worth getting an interface with more than one Out connection in case you add more units later. It is quite possible to control more than one MIDI instrument simultaneously with just one MIDI Out by holding them together in a 'daisy chain' sequence. However, if you add too many units in this way you may experience delayed reactions from those at the end of the chain. Giving each unit a chain lead from the interface will avoid this.

MIDI cables carry information on 16 independent channels, in effect allowing you to play up to 16 voices at once from any combination of your MIDI instruments. This limit can be exceeded with special MIDI interfaces such as Triple Play Plus which sports three sets of 16 channels.

So what constitutes a MIDI instrument? Technically it's anything that can send or receive MIDI data. Typical examples include controller keyboards, synthesiser keyboards, synthesiser modules and drum machines. Various other bits and pieces can also be controlled via MIDI, including many effects processors, some mixing desks and even lighting rigs for stage shows. Most MIDI instruments have no built-in speakers as they are designed to be routed through an optional mixer to

an amplifier and speaker system.

Only you can decide exactly what MIDI equipment you'll need to build your particular system or expansion. However, some pointers are always handy and to this end here are a few recommendations for anyone just starting out with a limited budget.

MIDI controller keyboard

MIDI controller keyboards look like synthesizers, i.e. make no sound of their own. The purpose of a MIDI controller keyboard is to transmit MIDI information to another unit, which would typically be a MIDI module or a sequencer.

The most basic controller keyboard would respond to and transmit both more than note-on and note-off information. In other words, which keys are pressed and for how long. However, you will respond to and transmit other information such as velocity (the force with which the keys are hit) and modulation data. Modulation is normally controlled by a wheel protruding from the left hand side of the keyboard and can be assigned to a particular aspect of the control sound. For example, if modulation was assigned to the sound's filter, moving the modulation wheel would have the effect of opening and closing the filter. A controller keyboard is only necessary if you have no other method of playing your MIDI instruments or your current MIDI keyboard is defective or restricted in features, as ours.

Synthesiser keyboard

One of the best ways to expand your Arranger is a MIDI system is with a synthesiser keyboard. This will act as a sound source and a controller keyboard. If this is to be your first purchase it's best to go for a mid-instrument keyboard. This will come with

a range of different sounds and will be able to play a number of them at the same time when used with your sequencer.

Most entry-level synth keyboards conform to the General MIDI (GM) standard. By definition, all GM instruments have the same set of sounds stored in the same locations. For example sound number one of any GM instrument is a piano. The set of GM sounds attempts to cover the range of 'lead' and 'batter' sounds that would be used in the production of a wide range of musical styles, including dance, rock, pop, orchestral and more. As you can imagine, one set of GM sounds is much like another (that is the whole point) which means that you are unlikely to find much individuality in a low-cost GM instrument.

However, there is nothing to stop a GM instrument having its own sounds and synthesis methods in addition to these, and there is also nothing to say that a keyboard synthesiser must comply to GM. You'll find entry-level keyboards that will have all the basic sounds you need to make a complete song, including drum sounds.

MIDI module

A MIDI module is a MIDI instrument that has no keyboard and can only be played or controlled by an external sequencer or other MIDI device. Like the previously discussed keyboard synthesizers, MIDI modules are often multitrack and GM compatible. Many synthesizers are also available in MIDI and GM format and are cheaper in comparison as you are not paying for the keyboard.

MIDI modules are handy for expanding a MIDI system when there is already a suitable controller keyboard in the set-up. If you were to use OctaMIDI or SoundBlaster it would be quite possible to run each straight up to the module with no need for a

MIDI keyboard, as these programs allow you to play the instruments from rather crudely from the Atmega keyboard. Most other sequencers do not have this function.

Drum machines

It's often a good idea to add a drum machine to handle your rhythm sections. Although many synths and modules come with drum sounds, sometimes they can be token exchanges and not really offer the kind of sound you want. Most drum machines come with pads which can be made to act like different drums, low for the bass drums, one for the snare so that you can beat out a rhythm in real time. Many are also equipped with their own sequencer which can be handy for testing out ideas even if you do have a sequencer running on your Atmega.

The only safe way to choose a drum machine is to try it out for yourself. Most modern models do a fair job of trying to please everyone. If you want to create an authentic live sounding rock back beat you will need to use a drum machine with sampled drum sounds which is the standard for most you'll find. However, if you really hate drums that shatter the rafters and all those weird sounds that crop up so often in electronic and dance music, an analogue machine is best for the job. Roland's TR808 and TR909 drum machines are responsible for these sounds and although they have been out of production for some years, many current digital drum machines have samples of these sounds. Some may try to capture the real others fail.

Whatever you do after, make sure you get in a good flesh around with any you are interested in before parting with your cash.

MIDI samplers

If you're hooked on samplers, the *Amiga* cannot compete with what the Amiga can offer in this, a lack of built-in MIDI samplers. Most come in the form of a MIDI module, although there are keyboard versions available. However, unlike modules, samplers have no internal sounds. Before you can get anything from a sampler you first have to either load in a sample from disk (or via SCSI or MIDI dump) or record a sound yourself.

The process of recording and editing on most MIDI samplers is a far cry from the on-screen graphical approach of Amiga sampling. This can be solved if you can find a way of transferring samples to your Amiga for editing, and then back to the sampler for playback. The advantage of a MIDI sampler is that you will normally get 16-bit resolution and higher sample rates, so the overall sound quality will be excellent. Also Roland and Kawai all produce a range of samplers, while the best entry level option is the Alesis 501, and the most discontinued units from Clavia.

Mixers

The essential function of a mixer is to combine a number of sounds from different sources into a single stereo signal. Mixers will also allow you to alter the stereo pan position, relative volume and total content (EQ) of the different sounds. A mixer is essential if your music system is made up of more than one instrument.

If you have a modest MIDI setup there's no need to splash out thousands of pounds on a professional studio mixing desk. You can buy a basic one or one with plenty of features for a home studio for as little as £159 - specifically the Yamaha Mix Pad # from Sound Technology. This offers three mono and three stereo line inputs (the mono inputs can be switched

to each with fader, bass and pan controls for each channel). Two auxiliary loops allow you to connect effects units and there's also balanced input and outputs.

Good quality entry-level mixers are also available from Steinberg Sound Technology and Mackie. Give them a call for details.

Effects processors

If you want a full sounding, rather subdued and lifeless, a probably tame in call in the help of an effects processor. Effects processors take an incoming sound and pass it through either analogue circuitry or digital algorithms to simulate effects such as echoes, reverb, distortion and so on. The affected sound is passed to the output. Vocals are nearly always swathed with reverb and various other effects, sometimes to gloss over imperfections in the voice.

An effects processor is not absolutely essential for producing professional quality music, but it will certainly help. The fact that you can apply many common effects directly to Amiga samples means that you can feel the liberty into thinking you have a bigger system than you actually have.

If you like to experiment and are on a tight budget it's worth taking a look at guitar effects pedals. These will work with any line level signal, not just guitars, and can be useful for adding a grungy warmth to sounds.

Recording

Recording straight to cassette tape is fine for demos but you'll need better quality for making master recordings. The DAT is now used by many MIDI musicians as a mastering medium (rather than the traditional multi-track reel-to-reel tape). DAT tapes (and some recorders) are easily portable and offer

CD quality recording

A cheaper option is to use DCC. This is a digital cassette tape but unlike DAT, the sound quality is not quite up to CD standard due to the compression system used by the recording and playback hardware. MiniDisc is on a par with DCC when it comes to recording quality. MiniDisc is like a tiny CD in a floppy disk-style case. However this is a more expensive option than DCC.

Finally there's hard disk recording, for which you could use your Amiga. This is covered in chapter two in the Digital Mastering section.

Glossary

The following glossary of terms includes short definitions used in this book to explain new terms. For example, the only other computer word that has in general use in the music industry

Amiga

Short for memory computer. Both hard and softwares often have to memory from which data is extracted for a low-level bus signal. As a consequence almost anything played on a Amiga that can be used for audio is low-level signal.

Balance (DALP)

Balanced audio connections are used for microphones and other signals where connection is likely to damage the signal. Two copies of the same signal are sent through the connection and the return copy of the other. There are two compared and any noise that has appeared in either signal is then canceled out.

BBPM

Acronym stands for Bit-Block-Play (BBP) is used to indicate the speed of the playback of a piece of music.

Broadcast

It simplest form is that of live broadcasting, often used as an instant delivery tracking track for shows records. The broadcast of all performances, normally cut up and packaged to advertise, is like this.

Character

Memory and data is divided by each audio input. Most characters allow for storage of more sounds. MIDI information that becomes data in its own character which are completely unconnected with sound playback.

DAT

Acronym: Digital Audio Tape. DAT is known from the standard form for a studio recording of a piece of music. DAT stores music by sampling it and writing the digital samples information onto magnetic tape. The mechanics of a DAT recorder are based upon VHS video technology with the addition of switches in digital and digital to analogue converters. A DAT cassette is smaller than a tape of magnetic and is used for music recording.

DCC

Acronym: Digital Compact Cassette. Like DAT, DCC records sound using frequency, not as digital information. The resulting recordings are for playback. It does not present digital analogue tape but not as high fidelity as those obtained by DAT. But it increases DCC into a technique to get more information in less time by jumping steps. Processing of the sound is very

levels are without significantly reducing the perceived fidelity. DCC players themselves can also play analogue cassette tapes but special DCC tapes are required for digital recordings.

dB

Decibels. Audio equipment measures the volume of a sound (magnitude of the signal) in decibels. The reference is 0 dB. (Zero dB is equal to 1 volt.)

Digital Processing

The enhancement or alteration of a sound. The most common use is the addition of effects and reverb which create a feeling of greater space in a sound. This can be current and in addition with reverb effects and/or applied to sample sounds from within multiple editing software.

DJ

Disc jockey. The term describes the relative levels of various frequencies in a sound. To DJ a sound is to alter the relative balance of its component frequencies (such as increasing the bass and decreasing the treble content for example).

Dolby

A noise control system in many decks. Peters usually after the volume control of a specific track.

Dolby

Quality describes the level of reproduction of an original sound. A high-quality (HQ) reproduction and sound almost as pure as pure as the original.

Dither

A component of a noise or synthesizer which contains specific frequencies often used as it passes through digital equipment. There are two key types in the creation of noise and dither.

Dolby DR / DTS

An extension to the Dolby system which allows that an enhancement must have certain points played at specific locations to allow a surround recording to use Dolby enhancement in only just the same or another Dolby environment.

Dolby Digital Processor

A system which records an audio signal in a number of track digital form in a surround format digital file. Complete tracks have their own separate processing tracks but these are mixed down into a surround track.

Dolby

Decibels. Literally 'levels per second' (Dolby) 'levels per second'. The amount to specify the sound and replay rate of a sample. For example a 44000 sample consists of 44 000 samples taken for every one second of 44 000 samples.

Dolby

Dolby plays are used to prevent illegal recordings, broadcasts and interceptions of other radio equipment and areas of their transmission. 2 hours and 4 hours. Both are available in stereo and mono versions.

DR

Acronym for generally described as one of two terms. Low and mid/LD/LD/LD signals are usually keyboards, guitars and other electric musical instruments.

Draconian Recordings

The best recording of a piece of music. Draconian duplicates are made on cassette, CD, vinyl or other media. It is vital that the quality of a track recording is as good as possible in order to ensure the acceptable loss of fidelity that is to be duplicated into analogue media such as cassette tape or vinyl.

DRMS

Digital Recording Management System. DRMS is a basic level of security. See LDRS.

DRM

Digital Rights Management Digital interface. This is the communication standard used by electronic musical instruments (MIDI) computers or musical equipment.

DRR

A device used to convert two or more sounds from different sources into a single signal, normally used to combine a number of instruments, one per stereo signal.

Dreher

1. A single module design available for the Mac system by Roland company. 2. MIDI module (MIDI module) an enhancement of a sequencer or any musical MIDI device that can receive and send data played via a computer using MIDI (most of them part).

Dribbles

A stereo sound is just a single signal, as opposed to a stereo signal with a left (usually slightly different) signal. When used to describe a synthesizer (sequencer) it is described the mono audio return to the left/right can be made to play more than one note at a time as opposed to a sequencer, which can play more than one note at a time, creating the playing of chords.

Dribbles

The traditional form of recording music on paper. Music sequencers have better form of storage devices to complement the traditional music form. See section of musical programming.

Dribbles

Any generated frequency waveform. The most popular waveforms of these are sine and saw.

Dribbles

A lot of them (over 100). If you want to sample a sound and you have the frequency range of that sound, the sample has to match the highest frequency present in the original sound.

Plane

The term used to describe the position of a sound within a stereo field. For example, a sound that was predominantly created from the left speaker would be referred to as being in the left.

Parallel Port

One of the original most used interface in the parallel port, which offers relatively small distinction of data between the Analog and digital interface. Because of this the parallel port is used by most Amiga computers due to their demand for very fast rates of data transfer.

Parrot

The title of the Amiga action track with music superimposed over it.

Parrot/PCA

Parrot and PCA structures are commonly used to convert CD players and other digital players to an amplifier. The Amiga's audio output is converted through a pair of parrot modules.

Parrot, Bob

The method used by musicians to display pieces of music on screen. A piano roll displays the horizontally scrolling sequence of notes, in which the position of each note is represented by a line, its pitch determined by its vertical position on the roll and its duration determined by the length of the line. Some piano roll machines will play the velocity or volume of each note with an additional method for each note.

PC/MOMA

Acronym: Personal Computer/Multimedia Computer Interface Adapter (or Personal Computer/Memory Card Interface Adapter) depending on who you believe. The Amiga 1200 and 8000 have been fitted with a PC/MOMA interface. This is found in the side and left hand ports of these Amigas and with the exception of some data, this offers the highest rate of data transfer for external peripherals. Due to this PC/MOMA is used by some Amigas and some of games.

Characteristics:

In the purpose of a MIDI sequencer in the simplest form, is playback. There may be several recorded notes from various instruments. Instruments occurs when a sequencer displays MIDI-packets showing the incoming signal + given number of notes, every second. If this note is too slow, it will not be apparent during playback, as the timing of each note will have been exaggerated to a reasonable degree.

Phasers

Surf Wave Processing

Phasers

Surf waves and switch ripples are options which can produce sound with phaser and feedback from memory. When the Amiga is used, often new data will still be entered in RAM after re-loading.

Sampling

Sampling is the process of converting an analogue signal into digital information. The process of the noise or sampled thousands of times and used to build a digital picture of the noise. The samples done is known as a sample and the waveform is what that Amiga is in the sample.

Sampling

Sampling techniques used to record and replay tracks from MIDI instruments. Sampling operations are more fast and far more basic and harder to use than complex based software techniques.

Sampling

Amiga samples are made up of two separate sounds, one for the right speaker to play and the other for the left.

Sampling

A device that creates musical sounds by use of samples of sounds. Samples samples are then then and then when the sound is passed it through synthesis and reprocessing, when when it produces the synthesis of sounds and enhance them to produce the more complex sounds.

Samples

Surf Wave Processing

Samples

Samples (collected). A type of memory card and memory card will give the user several over a variety of specific parameters.

Sampling

The sound of the rhythm. Sampling is generally specified in BPM (Beats Per Minute).

Sampling

Used in drum for the area concerned for a single part of a drum or note. Sampling, can also be known as drum pattern. This is often used as a term to describe when press of memory, usually one drum of visual parts.

Sampling

A type of sequencer attempt primarily designed for making music, with the Amiga's original sample-player features. Patterns are passed towards the Amiga's audio-only memory resources.

Sampling

The strength of a note. Velocity controls keyboards and drum pads are used to record the amount of pressure that is exerted on a key or pad as for and can affect the resulting sound accordingly. For example, striking a piano key with great or little force will be louder or louder on the key and it velocity control can be used with velocity. The Synthesizer often has velocity as a variable parameter of the sound, such as the filter frequency - in which case a higher press of a key would normally open the filter more.

Sampling

A component placed in other drives which changes the strength of a signal (volume of a sound). It creates normally reduces them of a synthesizer signal (volume control) up to its signal strength measured in dB.

Whitirea Models

A small range of standard frequencies (between 400 and 2000MHz) and many off-the-shelf synthesizers can often make a basis for synthesizer parts models.

Resonators

Allegro models as a two part system (oscillator and memory bank of configurations) are equipped with Zorro slots for the addition of expansion boards or memory and communication modules which allows fast and direct interchange between the Zorro slot for the expansion card and the larger RAM. There are three variants of the Zorro interface. Zorro 2 is the most memory and although this has been superseded by Zorro 3 which offers improved performance. If you have an 8088/80386 or 386SX memory board systems are available that convert your old Zorro 2 board to a Zorro 3.

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